

design ideas

Edited by Bill Travis and Anne Watson Swager

Dual-voltage supply powers SIM card

Larry Suppan, Maxim Integrated Products, Sunnyvale, CA

GLOBAL-SYSTEM-FOR-MOBILE-COM-
munication phones have a
subscriber-identification

module (SIM) that allows local wireless providers to recognize the user and his or her billing information. Although most SIMs are changing to 3V operation, they also accommodate 5V as well during the transition. IC₁ in **Figure 1** combines a step-up dc/dc converter with a linear regulator, allowing it to regulate up or down for a range of input voltages. It offers hardware-selectable fixed outputs of 3.3 and 5V; however, 3.3V is out of spec for a 3V SIM card. With properly chosen R₁/R₂/R₃ values, you can switch the regulated output between 3 and 5V (or any other two outputs within the allowed range) by applying digital control to the power-good input (PGI). The power-good output (PGO), the output of an internal comparator, then changes the IC's feedback by grounding the node between R₂ and R₃. If the power-good comparator is in use, you can implement the digital control using the 3/5 input and an external MOSFET (**Figure 2**). (DI #2468)

TO VOTE FOR THIS DESIGN,
CIRCLE NO. 315

Dual-voltage supply powers SIM card..... 113

Design formulas simplify classic
V/I converter..... 114

Rail-to-rail op amp provides
biasing in RF amp..... 114

Circuit multiplexes automotive sensors..... 116

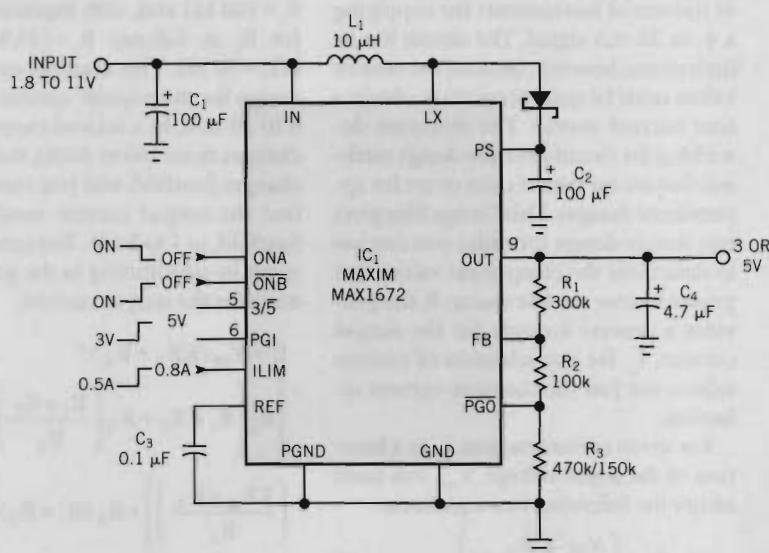
Analog switch acts as dc/dc converter..... 118

Circuit provides message
on disabled phone line..... 120

Optocoupler isolates shift registers..... 122

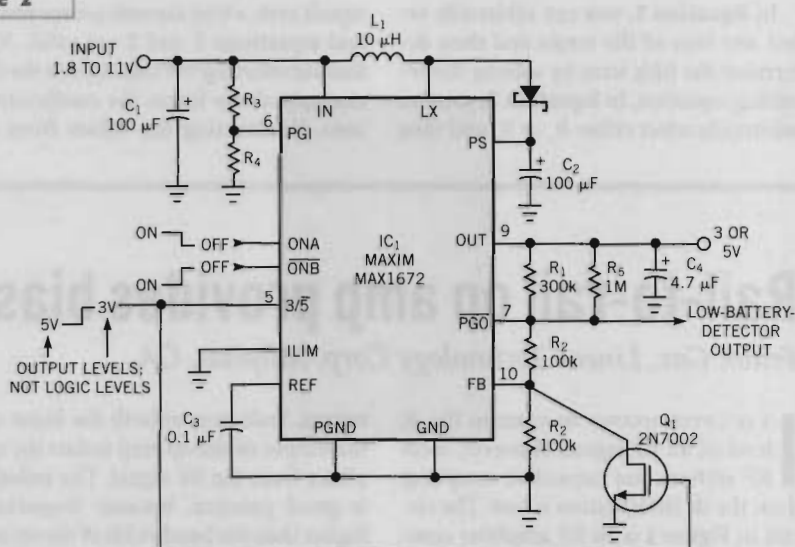
Tack a log taper onto a digital
potentiometer..... 124

Figure 1



You can obtain a regulated 3 or 5V output, according to digital control applied to the power-good input (PGI).

Figure 2



This circuit provides the same outputs as the circuit in Figure 1 without tying up the internal power-good comparator.

Design formulas simplify classic V/I converter

Dudley Nye, Nye Engineering Co, Fort Lauderdale, FL

FIGURE 1 SHOWS a classic voltage-to-current (V/I) converter. You can select the resistor values such that the output current in the load, I_L , varies only with the input voltage, V_{IN} , and is independent of R_L . The circuit is widely used in industrial instruments for supplying a 4- to 20-mA signal. The circuit has its limitations, however, because the resistor values must be quite accurate to obtain a true current source. The literature describing the circuit provides design methods that are for special cases or are for approximate designs. This Design Idea gives two simple design formulas you can use to determine the component values that produce a true current source. It also provides a general formula for the output current, I_L , for any selection of resistor values, not just the constant-current selection.

For a true current output, I_L , as a function of the input voltage, V_{IN} , you must satisfy the following two equations:

$$I_L = \left(\frac{V_{IN}}{R_1} \right) \left(\frac{R_2}{R_X} + 1 \right). \quad (1)$$

$$R_3 = (R_2 + R_X) \left(\frac{R_4}{R_1} \right). \quad (2)$$

In Equation 1, you can arbitrarily select any four of the terms and then determine the fifth term by solving the resulting equation. In Equation 2, you can arbitrarily select either R_3 or R_4 and then

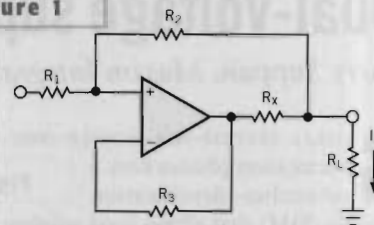
determine the unselected resistor after substituting the applicable terms from Equation 1. For example, you can solve Equation 1 for R_2 when $I_L = 20$ mA, $R_1 = 100$ k Ω , $R_X = 0.1$ k Ω , and $V_{IN} = 4$ V yields $R_2 = 49.9$ k Ω . Now, let $R_4 = 100$ k Ω and, with Equation 2, solve for R_3 as follows: $R_3 = (49.9$ k $\Omega + 0.1$ k $\Omega) = 50$ k Ω . This example configures a design for the popular current source of 4 to 20 mA. In a second example, if R_X changes from 100 to 400 Ω , the feedback changes fourfold, and you would expect that the output current would change fourfold, to 1 to 5 mA. You can check the result by substituting in the general formula for the output current:

$$I_L = V_{IN} (KR_2 + R_X) / \left\{ R_L \left[R_1 + R_2 + R_X \left(\frac{R_1 + R_2}{R_2} \right) - R_1 \cdot \left(\frac{KR_2 + R_X}{R_2} \right) \right] + R_X (R_1 + R_2) \right\}, \quad (3)$$

$$\text{where } K = 1 + \frac{R_3}{R_4}.$$

When the complete coefficient (the terms inside the square brackets) of R_L equals zero, a true current source results, and equations 1 and 2 are valid. Note that substituting the values from the first example above forces the coefficient to zero. Substituting the values from the

Figure 1



Design formulas make this classic V/I converter easy to use.

second example above results in the following expression:

$$I_L = 75.25 \frac{V_{IN}}{0.06R_L + 59.96}. \quad (4)$$

With $R_L = 0.2$ k Ω and $V_{IN} = 4$ V, $I_L = 5.019$ mA. Then, with $V_{IN} = 0.8$ V, $I_L = 1.003$ mA. Thus, after changing the feedback resistor by 4-to-1, you still have currents close to the 1- to 5-mA standard. Note also that $I_L = 5.02$ mA when $R_L = 0$ Ω ; thus, the circuit is still almost a perfect current source. This result is unique, as you can convert from 4 to 20 ma to 1 to 5 mA by changing only one resistor. You can configure the less used standard of 10 to 50 mA by making $R_X = 100/2.5 = 40$ Ω . (DI #2471)

TO VOTE FOR THIS DESIGN,
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Rail-to-rail op amp provides biasing in RF amp

Frank Cox, Linear Technology Corp, Milpitas, CA

IT IS OFTEN USEFUL to monitor the dc level of an RF signal. However, most RF systems use capacitive coupling; thus, the dc information is lost. The circuit in Figure 1 is an RF amplifier comprising two monolithic microwave integrated circuits (MMICs), IC_1 and IC_2 , and a quad rail-to-rail op amp (IC_3 , an LT1633). IC_{3A} restores the dc level at the

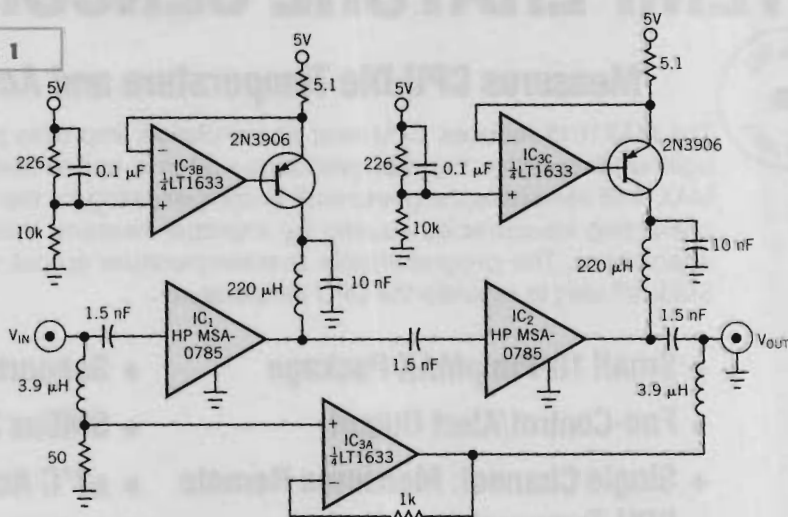
output. Inductors at both the input and the output of the op amp isolate the amplifier from the RF signal. The isolation is good practice, because frequencies higher than the bandwidth of the op amp can undergo rectification in the amplifier's input stages, thereby introducing offset. MMICs IC_1 and IC_2 are Hewlett-Packard HP MSA-0785 devices, which

have an inverting gain of 13 dB; the result is a total gain of approximately 26 dB and a noninverted signal. IC_1 and IC_2 have a 3-dB bandwidth of approximately 2 GHz. The 1.5-nF blocking capacitors set the low-frequency cutoff at 2 MHz.

IC_1 and IC_2 have a 1-dB compression point of 4 dBm, or 1 V p-p, into 50 Ω , allowing for an input level as high as 18 mV

rms. The maximum output current of IC_{3A}, typically 40 mA with a single 5V supply, limits the dc level on the output to 2V into 50Ω. The output saturation (low) voltage of the LT1633, typically 40 mV, sets the minimum pedestal voltage. IC₁ and IC₂ use constant-current bias sources to stabilize their gain with respect to temperature. Two other sections of the quad op amp, IC_{3B} and IC_{3C}, form active 22-mA current sources. You can make the voltage dividers on the noninverting inputs of IC_{3B} and IC_{3C} adjustable to trim the gain of the RF amplifier. The rail-to-rail inputs of IC₃ allow the circuit to operate to within 110 mV of the positive rail. (DI #2467)

Figure 1



A simple op-amp-follower circuit with the aid of inductive blocking restores the dc level of an RF signal.

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Circuit multiplexes automotive sensors

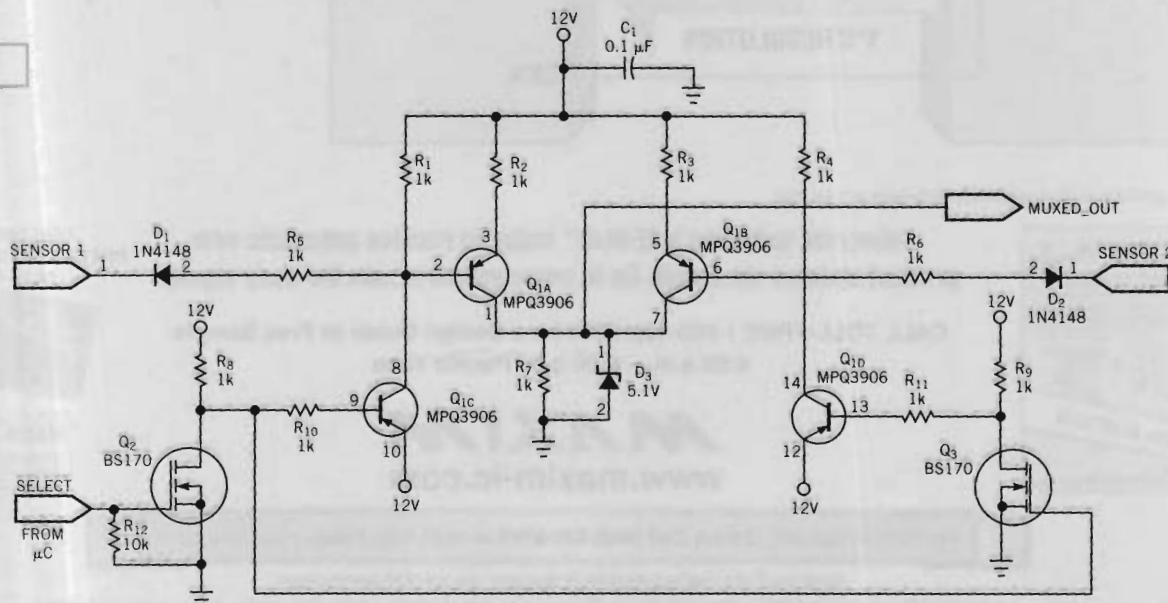
Adil Ansari, Delphi-Delco Electronics, Kokomo, IN

OFTEN, A μC limits the number of input-capture lines to accommodate the various types of automotive sensors with pulsed outputs, such as ve-

hicle- and engine-speed sensors. The circuit in **Figure 1** uses discrete components to multiplex two sensors with open-collector outputs into a single output, there-

by sharing one input-capture line of the μ C. The μ C selects the sensor whose output you will measure. You can apply this approach to sensors whose outputs are

Figure 1



You can multiplex the output signals from two sensors into one input-capture line in a μ C.

amenable to time-sharing and do not require continuous monitoring, such as position sensors. In **Figure 1**, Sensor 1 and Sensor 2 are outputs from two sensors using npn transistors with open-collector outputs. To enable Sensor 1 or Sensor 2, Q_{1A} or Q_{1B} , respectively, must turn on. A logic-low signal from the μC on the Select input turns off Q_2 and Q_{1C} . When Sensor 1 input goes low, D_1 forward-biases, and Q_{1A} turns on, providing a high signal on MUXED_OUT. When Sensor 1 input turns off (high-impedance state), Q_{1A} turns off, providing a low signal on

MUXED_OUT. Therefore, when the Select input is low, MUXED_OUT produces pulses that are inverted but synchronized with the Sensor 1 pulses. At the same time, Q_3 and Q_{1D} are on, turning off Q_{1B} and disabling the Sensor 2 input.

Similarly, when Select goes high, Q_2 and Q_{1C} turn on, turning off Q_{1A} and disabling the Sensor 1 input. At the same time, Q_3 and Q_{1D} turn off, allowing the Sensor 2 signal to turn Q_{1B} on and off when Sensor 2 switches on (low) and off (high-impedance state), respectively. Therefore, MUXED_OUT produce puls-

es synchronized with the Sensor 2 input. You can change the values of R_1 , R_4 , R_5 , and R_6 to meet the sensors' requirements. D_3 clamps MUXED_OUT to CMOS/TTL levels. The use of the MPQ3906, containing four pnp transistors in one package, minimizes the number of components. Similarly, you can obtain arrays of 1-k Ω resistors in a single package. (DI #2469)

TO VOTE FOR THIS DESIGN,
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Analog switch acts as dc/dc converter

John P Skurla, Advanced Linear Devices Inc, Sunnyvale, CA

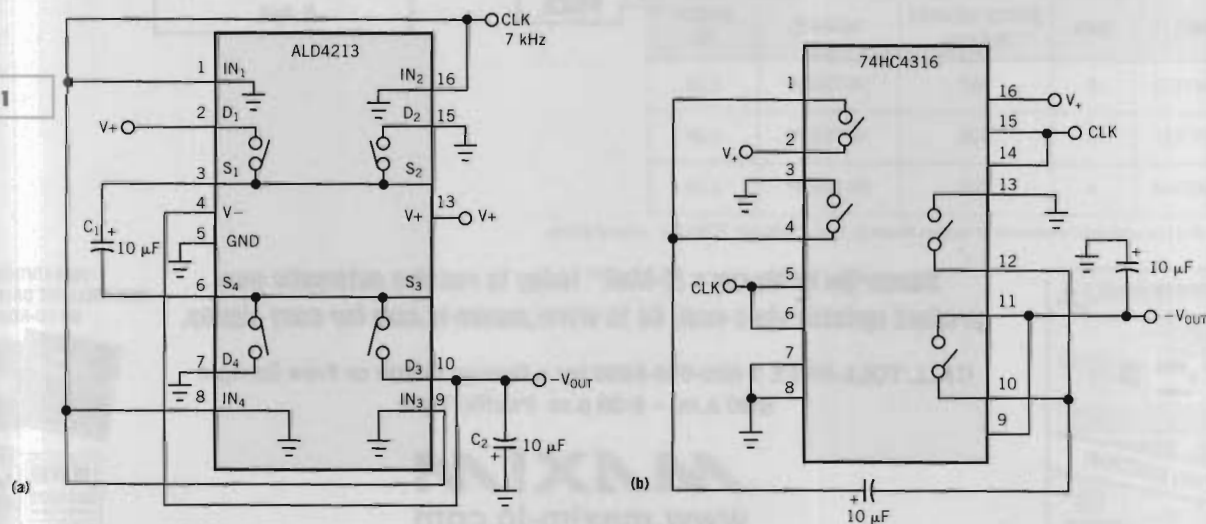
MANY LOW-CURRENT DEVICES that require 65V supplies can operate reliably in a single 5V power-supply environment if you use an appropriate localized dc/dc converter to generate the -5V bias. Often, the capabilities and advantages of these 5V ICs far outweigh the minor inconvenience and added costs

of an additional -5V-converter function. Many companies manufacture dc/dc-converter ICs and modules in a variety of power ratings and footprints. However, these typical dc/dc converters can be overkill for simple, single-chip applications that require only a negative bias voltage with low operating currents. For

these applications, typical negative-voltage requirements range from -4 to -6V with a supply current of 1 mA, and requirements for the -5V supply are generally noncritical.

A lower cost alternative to conventional dc/dc converter modules for generating negative dc voltages from a posi-

Figure 1



NOTES:
 $2V \leq V_+ \leq 5V$.
 CLK IS CMOS LOGIC LEVEL WITH FREQUENCY OF 5 TO 500 kHz.
 V_+ IS THE DC-TO-DC INPUT.
 $-V_{OUT}$ IS THE DC-TO-DC OUTPUT.

Using an analog switch with two external capacitors and an external clock is a viable way to produce 25V from a 5V input for low-power, -5V needs. One approach uses only one phase of the clock (a); a second approach requires both phases (b).

tive supply uses a low-cost quad-semiconductor analog switch and an onboard system clock (Figure 1a). This type of voltage converter generates a low-power, negative bias voltage from a 5V input. This circuit emulates charge-pump dc/dc converters, which are suitable for generating an output voltage whose polarity is opposite that of the input voltage. Two charge-storage capacitors are also necessary, as with conventional converters. Unlike the conventional self-contained dc/dc converter approach, this circuit requires a single external clock input to sequence the switches on and off and approximately the same amount of pc-board space. You can tap this clock from any 5V logic-gate output with continuous, regular periods of 5- to 500-kHz signals.

Charge-pump converters operate by first charging up one capacitor and alternately transferring that charge to another capacitor using a switching circuit. The switching circuit in Figure 1a alternately charges and discharges C_1 and C_2 to generate a -5V output from a 5V input. Integrated level translators and logic

gates inside the ALD4213 analog switch provide the logic translation to convert a single 5V input to a $\pm 5V$ logic swing.

The circuit closes two switches, S_1 and S_4 , under clock control. During the first half of a clock cycle, C_1 charges up to a voltage equal to the input voltage, $V+$. The next half-cycle of the clock control opens S_1 and S_4 and closes S_2 and S_3 . C_1 now connects across C_2 through S_2 and S_3 , and the charge on C_1 subsequently transfers to C_2 until the voltage across both C_1 and C_2 is equal. Notice the "inverted" polarity across C_2 , which forces the output voltage on C_2 to be $V-$, or the opposite of $V+$.

Each subsequent clock cycle, which again begins with the closing of S_1 and S_4 , causes C_1 to charge up from the previous voltage to $V+$. After many repeated clock cycles, the voltage on C_2 remains charged to a value equal to the negative of $V+$, or close to it; it performs the function of a voltage inverter, which is more commonly called a converter.

An alternative analog-switch-based converter uses the industry-standard

74HC4316 quad analog switch with level translator (Figure 1b). The circuit is similar to the circuit in Figure 1a but has different pin connections. This circuit also requires both phases of the clock. You can use an additional inverting logic gate to generate both clock phases if necessary. The recommended input is a logic clock that has a useful frequency range of 5 to 500 kHz.

Figure 1a's single-phase design costs less than \$1 in large quantities. The cost of the circuit in Figure 1b can be less than half the cost of the circuit in Figure 1a provided that both clock phases exist and that you don't have to add an external logic-gate inverter. You can also integrate analog-switching inverters with other analog functions in a custom ASIC; the ALD4213 and ALD500A are compatible with the company's library of standard cells. (DI #2476)

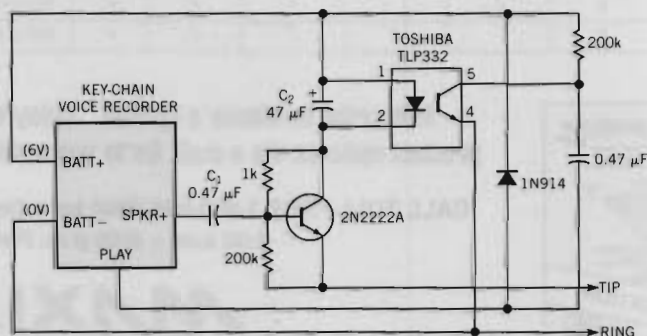
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Circuit provides message on disabled phone line

Kevin Kelley, BAE Systems, Greenlawn, NY

PHONE COMPANIES OFTEN disconnect a misbehaving phone line from a complainant's residence for troubleshooting purposes. With the problem between the residence and the central office, the residence is left with a dead phone line and no visible repairman while the line is under repair. The circuit in Figure 1 adapts a small key-chain voice recorder to the Tip and Ring lines of a phone line that has been disconnected from the central office. The purpose is to play a prerecorded message into any phone on the line when its receiver goes off-hook. A Radio Shack key-chain voice-memo recorder (part number 63-945) or a similar device provides solid-state voice-message storage and playback in a small package and also

Figure 1



Dead phone line? You can send a prerecorded message to any phone on the defunct line.

powers the phone line with its internal 6V batteries. You open and modify the recorder to bring four signals out to the external circuit: Battery (+), Battery

(-), Speaker (+ or -), and the Play button contact. You can disconnect the internal speaker to save power.

With all phones on-hook, phone-line

current is near zero, keeping the optocoupler off and its transistor open with the voltage at Pin 5 at the battery voltage. The Tip and Ring lines are at 6 and 0V, respectively, to power the phones on the line (Most phones operate on as little as 3V.) Battery drain in this condition is minimal. When a receiver goes off-hook, the line impedance drops, and several milliamps flow through the saturated

transistor. The transistor provides a high ac impedance between C_1 and the battery, allowing audio-signal transfer to the line, and provides a low dc resistance to maximize the low battery voltage to the phones. The transistor current turns the optocoupler on, and the voltage on Pin 5 drops to near 0V. This negative edge generates a low pulse into the Play contact, as if you had pressed the Play but-

ton. The message plays once in its entirety every time a receiver goes off-hook. C_2 prevents any clicks at the end of the message from restarting the sequence if the receiver goes on-hook before the message ends. (DI #2472)

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Optocoupler isolates shift registers

Jim Hartmann, Silent Knight LLC, Maple Grove, MN

CONVENTIONAL SHIFT REGISTERS, such as the 74HC595, require data-, clock-, and strobe-logic signals. The circuit in Figure 1 needs only two logic signals to isolate and control shift-register devices. For each transmitted bit and one of the two optocouplers receives a short drive pulse: one optocoupler for a high transmitted bit and the other for a low bit and After pulsing all the bits, the circuit a final concurrent 1 and 0 pulse strobes the data into the output registers. Two logic-gate packages on the isolated side of the circuit decode the two negative pulse signals back into data, clock,

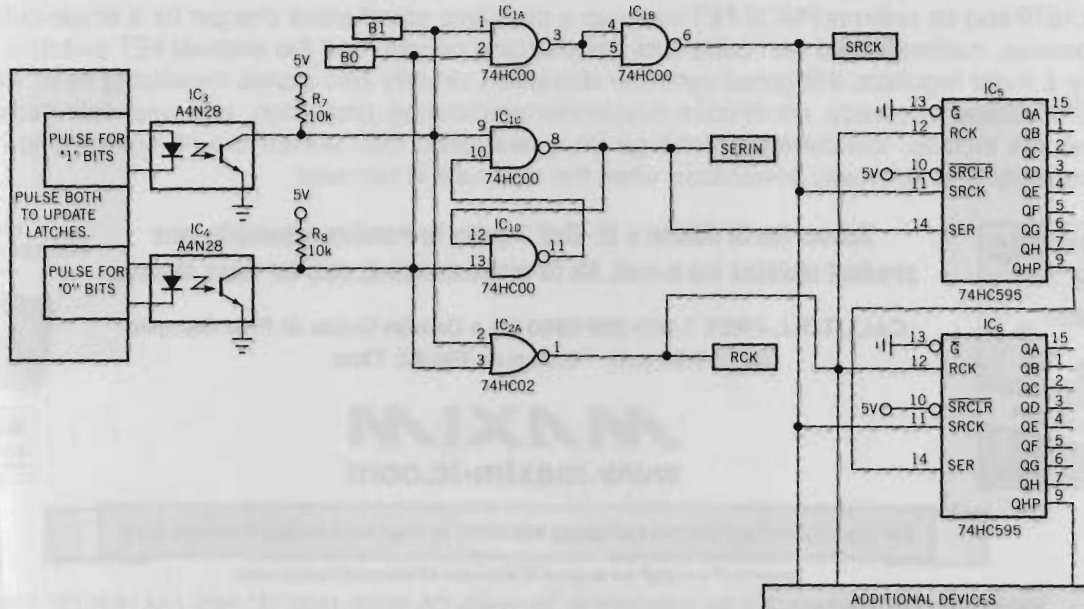
and strobe. Two NAND gates form an RS latch that captures the data state for the serial input (SERIN). Two more NAND gates form an AND to combine the two pulse sources into the SRCK shift clock. Finally, a NOR gate (or four more NAND gates) produces the RCK strobe. You can cascade the shift-register devices as necessary.

You have no timing constraints on the signals other than observing the maximum data rate of the optocouplers and ensuring an off period between pulses. The final latch pulse also generates an extra rising SRCK edge that you can use to

load the first bit of the next sequence. In this case, the optocoupler that turns off last determines the RS latch state for the first bit. You can also ignore the extra clock; it has no effect on the output. Low power consumption is possible by keeping the pulses as short as possible by limiting the LED current and the updating rate. For example, with 40- μ sec pulses and 1-msec period, the average drive current is 80 μ A. (DI #2470)

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Figure 1



Optocouplers allow you to isolate and control shift registers with only two logic signals.

Tack a log taper onto a digital potentiometer

Hank Zumbahlen, Analog Devices, Campbell, CA

IT'S SOMETIMES CONVENIENT to have digital control of the volume level in an audio system. The use of multiplying DACs (MDACs) is problematic because of the switching noise of the ladder network. This noise comes from the bit switches injecting charge into the signal when they turn on and off. Audio engineers have dubbed this noise "zipper noise" from the sound that results from dynamically adjusting the volume (gain riding). An alternative to an MDAC in this application is a digital potentiometer, such as the Analog Devices AD52XX, AD84XX, or AD7376. You can think of the digital potentiometer as a tapped resistor string. It generates less noise because fewer switches change state. In addition, you can connect the three terminals of the potentiometer anywhere within the common-mode range of the circuit (the supply-voltage range), unlike an MDAC, which generally uses ground as reference.

The primary drawback with using the digital potentiometer for volume control is that it currently comes with only a linear taper. With a linear taper, if the "wiper" is at the midpoint, the signal is only 6 dB less than the maximum. Thus, most of the adjustment range occurs

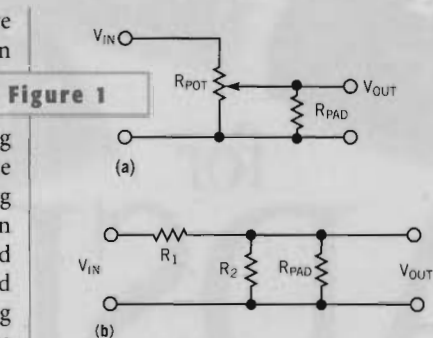


Figure 1
Adding a pad resistor to a digital potentiometer imparts a logarithmic-like taper to the device.

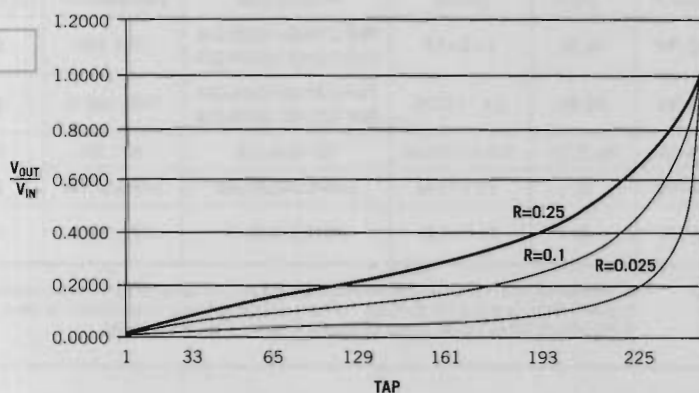
within a small percentage of the range of the potentiometer. This constraint limits the adjustability of the volume setting. The ear responds logarithmically; the volume control should respond similarly. The primary reason for having only a linear taper is the manufacturing problems that the large range of resistance values for a log taper cause. By adding a pad resistor from the wiper of the potentiometer to one end (Figure 1a), you can simulate a log taper. If you split the potentiometer into two resistors, R_1 and R_2 , you can redraw the circuit as in Figure 1b. The output voltage then depends on

the parallel combination of R_2 and R_{PAD} . You define a ratio, r , which is R_{POT}/R_{PAD} ($R_{POT} = R_1 + R_2$). By adjusting the value of R_{PAD} , you can modify r , which adjusts the taper, or the attenuation-versus-digital-input code to suit the application. The following expression gives the transfer function of the potentiometer:

$$\frac{V_{OUT}}{V_{IN}} = \frac{R_2 \parallel R_{PAD}}{R_1 + R_2 \parallel R_{PAD}}$$

Figure 2 shows the attenuation curves for three values of a pad resistor. As you can see, this trick doesn't give a taper that is so many decibels per step, but it does allow for better low-level settability. You must address a couple of issues. The first is that the end-to-end resistance of the potentiometer changes with the digital code. It varies from the potentiometer resistance at one end (with the wiper at the lower end) to the value of the pad resistance in parallel with the potentiometer resistance at the other end. If you configure the circuit as a typical attenuator and drive it from a low-impedance source, the low pad resistance should not present a major problem. If, however, you are trying to obtain a set resistance value to determine a time constant (or any other application in which the resistor value is critical), this approach may not work well. The second issue involves overvoltage. The three terminals of the potentiometer can be anywhere within the supply range of the IC, which is 5V for the AD52XX and $\pm 15V$ for the AD72XX family. If you apply overvoltage to one of the pins, even in a transient condition, the IC could latch up because of a parasitic substrate SCR. (DI #2473)

Figure 2



It's not log, but it's close. These curves approximate what you can obtain from an audio-taper potentiometer.

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